

Amendments to the Specification:

Page 4 lines 26-29 through Page 5, lines 1-12, replace this paragraph with the following amended paragraph:

In another embodiment, a method of the present invention involves the communication between a calling-from-party and a calling-to-party. The method includes the step of transmitting a uniform resource locator to the calling-from-party in response to an initiation signal transmitted from the calling-from-party. The uniform resource locator may be transmitted over a data link (e.g., data session). The uniform resource locator identifies a location where multimedia content may be accessed by the calling-from-party if the calling-to-party is a service subscriber according to a database of service subscribers. Subsequently, another data link (e.g., data session) may be established for the downlink of the multimedia content. A voice link may then be established to the calling-to-party in response to the initiation signal from the calling-from-party. Thereafter, the initiation signal may be bridged to establish a voice link between the calling-from-party and the calling-to-party. This bridging may rely on the previously established voice link. In response to establishing a voice link between the calling-from-party and the calling-to-party, the established data links (e.g., data sessions) may be terminated.

Page 5, lines 26-29, replace the two paragraphs with the two amended paragraphs as follows:

FIG. 5 depicts a flow chart of yet another embodiment of the present invention; and

FIG. 6 depicts a flow chart of still another embodiment of the present invention[.]; and

Page 5, after line 29, add the following paragraph:

FIG. 7 depicts a flow chart of still yet another embodiment of the present invention.

Page 8, lines 20-25, replace paragraph with the following amended paragraph:

Referring to FIG. 2, a flow chart depicting a further embodiment of the present invention is illustrated. Here, another algorithmic method 200 is shown for use with multiple network types. Method ~~100~~ 200 also depicts a sequence of steps in providing a collaboration of voice service(s), carried over a circuit-switched ("CS") type network, with multimedia service(s), carried over a packet-switched ("PS") type network.

Page 8, lines 26-29 through Page 9, lines 1-7, replace paragraph with the following amended paragraph:

Algorithmic method 200 initially provides for the step of transmitting an initiation signal from a calling-from-party (step ~~110~~ 210). This initiation signal is received by a communication network infrastructure element, such as a service control point/service node (e.g., server), for example. The step of transmitting the initiation signal may include entering the telephone number of the calling-to-party into the temporary memory of the calling-from-party's wireless or wireline unit. Thereafter, a corresponding send button or the like on the calling-from-party's wireless or wireline unit may be depressing so as to commence the sequence of steps in formulating a telephonic connection between the calling-from- and calling-to- parties. Consequently, the initiation signal identifies the calling-to-party.

Page 12, lines 6-25, replace paragraph with the following amended paragraph:

In one example of the call flow, a calling-from-party 340 may initiate a voice call to a calling-to-party 370 over circuit-switched network 310. In turn, a Mobile Switching Center ("MSC") may then ~~may~~ send the call control over to service control point/service node 330. Thereafter, service control point/service

node 330 may determine if the calling-to-party 370 is a subscribers to the multimedia service (e.g., video greeting service). If so, service control point/service node 330 may transmit a video clip URL to the calling-from-party 340. This step may be realized through a SIP INVITE. Once the video clip URL is transmitted, calling-from-party 340 may receive the multimedia content associated with the URL from video server 360. Video server 350, as a result, may send the multimedia content (e.g., video greeting) to the calling-from-party 340. Subsequently, service control point/service node 330 may initiate a voice call to calling-to-party 370. ~~After calling-to-party 370 the voice call~~ the voice call to calling-to-party 370 is established from service control point/service node 330, service control point/service node 330 may transmit a termination message (e.g., BYE) to calling-from-party 340 to terminate the SIP session in which the video clip may be transmitted. Thereafter, service control point/service node 330 may then bridge the two voice call legs together. Consequently, calling-from-party 340 and calling-to-party 370 may now have a complete voice conversation.